

SPECIFICATION

TITLE

METHOD FOR THE OPERATION OF A HEARING AID DEVICE OR HEARING
DEVICE SYSTEM AS WELL AS HEARING AID DEVICE OR HEARING DEVICE
SYSTEM

BACKGROUND OF THE INVENTION

FIELD OF THE INVENTION

[0001] The invention is directed to a method for operating a hearing aid device or hearing device system having at least one first microphone for generating a first microphone signal and a second microphone distanced from the first for generating a second microphone signal. The invention is also directed to a hearing aid device or hearing device system for implementing the method.

DESCRIPTION OF THE RELATED ART

[0002] Acoustic feedback frequently occurs in hearing aid devices, particularly for hearing aid devices having a high gain. This feedback is expressed in pronounced feedback-caused oscillations having a specific frequency, called "whistling", that is usually extremely unpleasant both for the hearing aid user and as well as for people in the immediate surroundings.

[0003] Feedback can occur when sound that is picked up via the microphone of the hearing aid device is amplified by a signal amplifier and output via the earphone proceeds back to the microphone and is re-amplified. In order for the typical "whistling" to occur (usually at a dominant frequency), however, two further conditions must be met. First, the "loop amplification" of the system, i.e., the product of the hearing aid gain and the attenuation of the feedback path, must be greater than 1. Second, the phase shift of this loop amplification must correspond to an arbitrary whole-numbered multiple of 360° .

[0004] The simplest approach for reducing feedback-conditioned oscillations is to permanently reduce the hearing aid gain so that the loop amplification remains below the critical limit value even in unfavorable situations. The critical disadvantage of this approach, however, is that the hearing aid gain required given more

Figure 1 consists of 12 histograms arranged in a 6x2 grid. The columns are labeled 'n = 10' and 'n = 20'. The rows are labeled 'm = 10', 'm = 20', 'm = 30', 'm = 40', 'm = 50', and 'm = 60'. The x-axis for all histograms is 'Number of non-zero elements' ranging from 0 to 100. The y-axis is 'Frequency' ranging from 0 to 10. The distributions are roughly bell-shaped and centered around 50 for n=10 and around 100 for n=20. As m increases, the distributions become narrower and taller.

Date of Deposit: March 1, 2002

BOX Patent Application
Assistant Commissioner for Patents
Washington DC 20231


Title: Method for the Operation of a Hearing Aid Device or Hearing Device System as well as Hearing Aid Device or Hearing Device System

Specification, claims and abstract;

Two sheets of drawings, Figures 1-2;

UNSIGNED DECLARATION;

Filing Fee
Postcard

Postcard

 Signature of person mailing documents and fees

pronounced hearing impairment can no longer be achieved as a result of this limitation.

[0005] Other approaches provide a measurement of the loop amplification during the hearing aid adaptation and reduce the hearing aid gain in designational fashion in the critical range with the assistance of what are referred to as notch filters (narrow-band blocking filters). Since the loop amplification, however, can constantly change during daily life, the benefit is likewise limited.

[0006] A number of adaptive algorithms have been proposed for dynamic reduction of feedback-conditioned oscillations, these automatically adjusting to the respective feedback situation and implementing corresponding measures. These methods can be roughly divided into two categories.

[0007] The first category comprises "compensation algorithms" that estimate the feedback part in the microphone signal with the assistance of adaptive filters and neutralize the feedback by subtraction and, thus, do not deteriorate the hearing aid gain. However, these compensation methods assume uncorrelated, i.e., ideally, "white", input signals. Tonal input signals that always exhibit a high time correlation lead to an incorrect estimate of the feedback path, which can lead to the fact that the tonal input signal itself is erroneously subtracted.

[0008] The second class contains the algorithms that only become active when feedback-conditioned oscillations are present. They generally contain a mechanism for detecting oscillations that continuously monitors the microphone signal. When feedback-typical oscillations are detected, the hearing aid gain is reduced to such an extent that the loop amplification drops below the critical limit. The gain reduction can ensue, for example, by lowering a frequency channel or by activating a suitable narrow-band stop filter (notch filter). This is disadvantageous because the oscillation detectors can fundamentally not distinguish between tonal input signals and feedback whistling. The result is that tonal input signals are interpreted as feedback oscillations and are then incorrectly reduced in level due to the reduction mechanism (for example, notch filters).

[0009] Delay elements that have a decorrelating effect are often introduced into the signal processing chain in the compensation algorithms in order to prevent tonal signal segments having a length characteristic for voice signals from being

hearing device system that avoids feedback-conditioned oscillations without noticeably deteriorating the sound quality.

[0015] This object is inventively achieved by a method for operating a hearing aid device or hearing device system, comprising generating a first microphone signal from at least one first microphone; generating a second microphone signal from at least one second microphone that is distanced from the at least one first microphone; comparing the first microphone signal and the second microphone signal; recognizing feedback-conditioned oscillations based on the comparing; and reducing the feedback-conditioned oscillations when they are recognized as such.

[0016] This object is also achieved by a hearing aid device or hearing device system, comprising at least one first microphone configured to generate a first microphone signal; at least one second microphone distanced the first microphone configured to generate a second microphone signal; a signal processing unit configured to process the first microphone signal and the second microphone signal; a comparison unit configured to compare the first and second microphone signals or signals derived from them and to recognize feedback-conditioned oscillations; and a feedback-conditioned oscillation reducer.

[0017] The invention can be employed in all standard types of hearing aid devices, for example, given hearing aid devices to be worn behind the ear, hearing aid devices to be worn in the ear, implantable hearing aid devices or pocket devices. Furthermore, a hearing device system composed of a plurality of devices can also be utilized for serving a hearing-impaired person, for example, a hearing device system having two hearing devices worn at the head for binaural coverage. The microphone signals that are analyzed for the recognition of feedback-conditioned oscillations can then also proceed from different devices.

[0018] In the invention, microphone signals of at least two microphones distanced from one another are generated. At least one microphone must be arranged such that it does not pick up feedback-conditioned oscillations of a hearing aid device or at most picks them up in highly attenuated form. Useful signals, however, should be picked up by the appertaining microphones in a similar way. By analysis and comparison of the microphone signals or signals derived from them, a distinction can be made between feedback-conditioned oscillations and useful

conditioned oscillations in the individual hearing aid devices occur at different frequencies. A tonal useful signal (for example, a sine signal), in contrast, appears at the same frequency at both sides. When an oscillation is detected at one side, then it is a feedback signal issue only when no oscillation at this frequency is detected from the microphone signal of the other hearing aid device. When, in contrast, an oscillation at the same frequency is detected at both hearing aid devices, then there is a great probability that this is a sine-shaped input signal.

[0022] In one embodiment of the invention, a correlation analysis is undertaken for comparing the microphone signals of two distanced microphones for recognizing feedback-conditioned oscillations. Different frequencies of feedback-conditioned oscillations in two microphone signals mean that no significant, correlated signal parts exist in the respectively other microphone signal for the oscillation signal of the one microphone. In the feedback case, the two microphone signals are thus only slightly correlated. In contrast, a high correlation is present in the case of a useful tonal signal. This is true not only of tonal signals; each signal coming from a useful sound source enters two hearing aid microphones distanced from one another with a high cross-correlation value.

[0023] When feedback-condition oscillations have been recognized from the comparison of microphone signals or signals derived from them, then reducing the hearing aid gain provides one possibility of suppressing these oscillations. When the signal processing in a hearing aid occurs in a plurality of parallel channels of a signal processing unit, then the hearing aid gain in one embodiment of the invention can be reduced only in the frequency channels in which feedback-conditioned oscillations are present.

[0024] The invention provides another possibility for reducing feedback-conditioned oscillations by eliminating these oscillations with narrow-band filters whose limit frequencies approximately coincide with the oscillation frequencies or with other feedback-conditioned oscillation reducers. For example, the filters can be implemented as notch filters. When one notch filter does not suffice, then further notch filters at the respective frequency are activated given a renewed detection of oscillations.

[0025] In another embodiment of the invention, when an adaptive filter for reducing feedback-conditioned oscillations is employed in a hearing aid device, then the adaptive compensation filter is adapted when feedback-conditioned oscillations are recognized. For example, the operating parameters of the filter can be varied such that the adaptation speed is increased. Conversely, the adaptation speed of the adaptive compensation filter is reduced when no feedback-conditioned oscillations are detected. This principle can be analogously applied to compensation filters based on the frequency range. Both the correlation analysis for recognizing feedback-conditioned oscillations as well as the regulation of the adaptation speed can advantageously take place in a frequency-specific manner.

[0026] When a hearing aid device of the invention recognizes feedback-conditioned oscillations on the basis of a correlation analysis of two microphone signals (cross-correlation), then there is a further possibility for reducing these oscillations by suppressing uncorrelated frequency parts of the microphone signals. Only those signal parts that are essentially uniformly present in all microphone signals are then further-processed.

DESCRIPTION OF THE DRAWINGS

[0027] Further details of the invention are explained in greater detail below on the basis of the exemplary embodiments shown in the drawings.

[0028] Figure 1 is a schematic block diagram of a hearing aid device in which feedback-conditioned oscillations are recognized by comparing two microphone signals; and

[0029] Figure 2 is a schematic block diagram of two hearing aid devices between which a signal exchange is provided for recognizing feedback-conditioned oscillations.

DETAILED DESCRIPTION OF THE INVENTION

[0030] The hearing aid device schematically shown in Figure 1 comprises a microphone 1, a signal processing unit 2 as well as an earphone 3. When sound from the earphone 3 proceeds back to the microphone 1, then feedback-conditioned oscillations (feedback) can arise. The conditions for this are that the "loop amplification" of the system, i.e., the product of the hearing aid gain and the attenuation of the feedback path, is greater than 1, and that the phase shift of this

[0031] As shown in Figure 1, the microphone signals of the microphones 1 and 4 can also be initially supplied to a respective signal pre-processing unit 6 and 7. The signal pre-processing can comprise, for example, A/D conversion or a signal pre-amplification.

[0032] Figure 2 shows two hearing aid devices 11 and 11' each having a microphone 12, 12', a signal processing unit 13, 13' and an earphone 14, 14'. Respectively one oscillation detector 15, 15' monitors the microphone signal continuously for oscillations and identifies the oscillation frequencies when oscillations are detected. A signal path 17 for the signal exchange between the hearing aid devices exists between the hearing aid devices 11 and 11'; this can ensue wirelessly or wire-bound. According to the invention, an exchange of the detected oscillation frequencies ensues via the signal path 17. The oscillation

